

# (12) UK Patent Application (19) GB (11) 2 312 135 (13) A

(43) Date of A Publication 15.10.1997

(21) Application No 9607571.8

(22) Date of Filing 12.04.1996

(71) Applicant(s)

**Lightworks Editing Systems Limited**

**(Incorporated in the United Kingdom)**

**Ariel House, 74A Charlotte Street, LONDON,  
W1P 1LR, United Kingdom**

(72) Inventor(s)

**Hugh Steers**

(74) Agent and/or Address for Service

**Langner Parry**

**High Holborn House, 52-54 High Holborn, LONDON,  
WC1V 6RR, United Kingdom**

(51) INT CL<sup>6</sup>

**H04H 7/00**

(52) UK CL (Edition O )

**H4R RSX**

(56) Documents Cited

**GB 2306750 A GB 2258374 A US 4479240 A**

(58) Field of Search

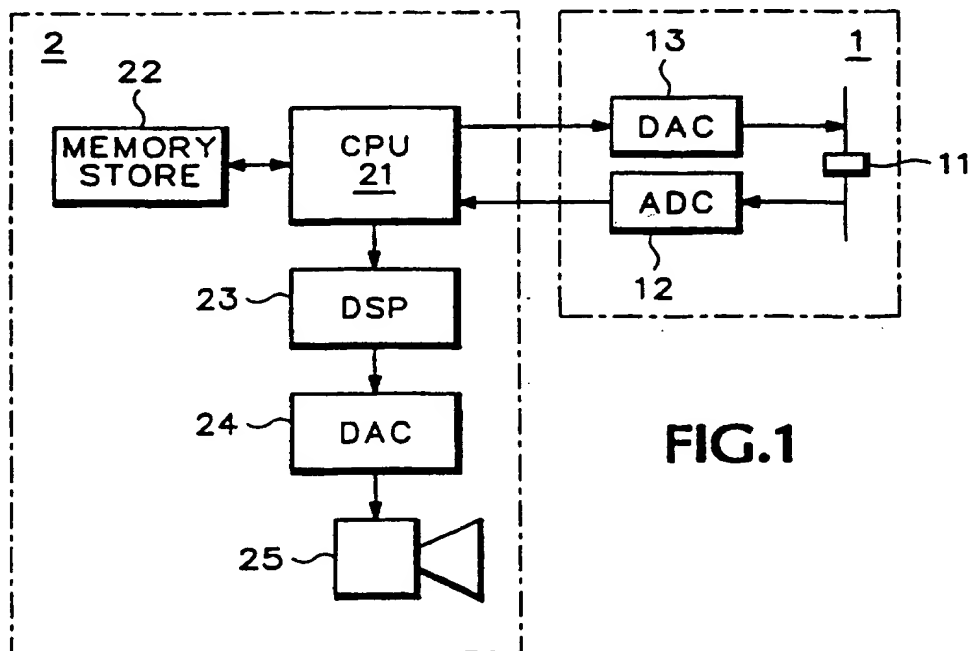
**UK CL (Edition O ) G5R RB81 , H4R RPNR RPX RSCB  
RSX**

**INT CL<sup>6</sup> G11B , H04H**

**Online: WPI, JAPIO, INSPEC**

## (54) Audio signal control system and method therefor

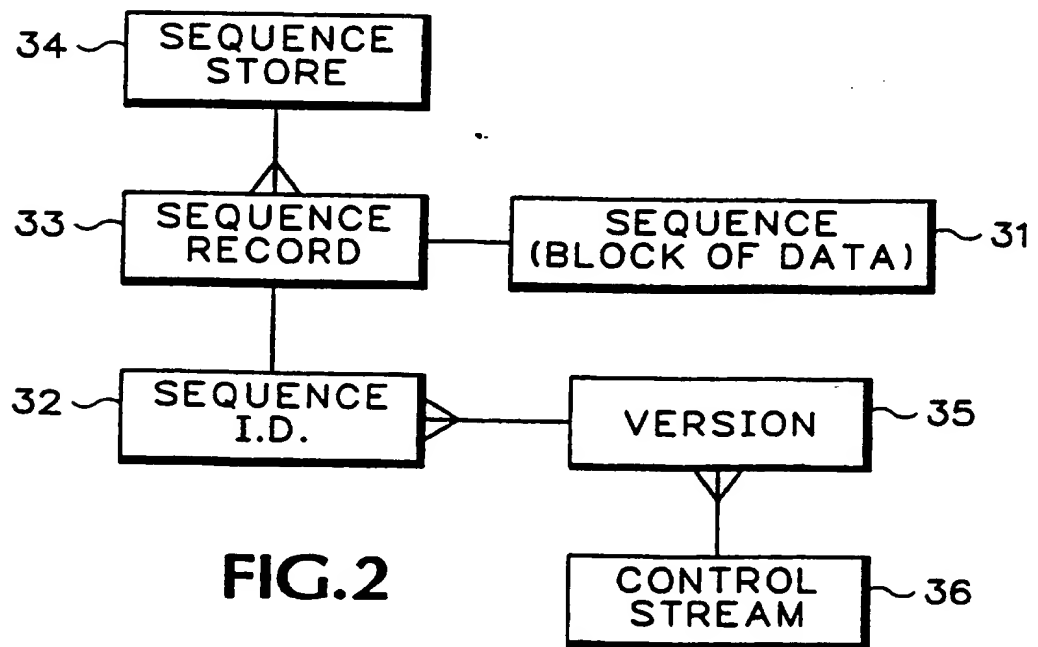
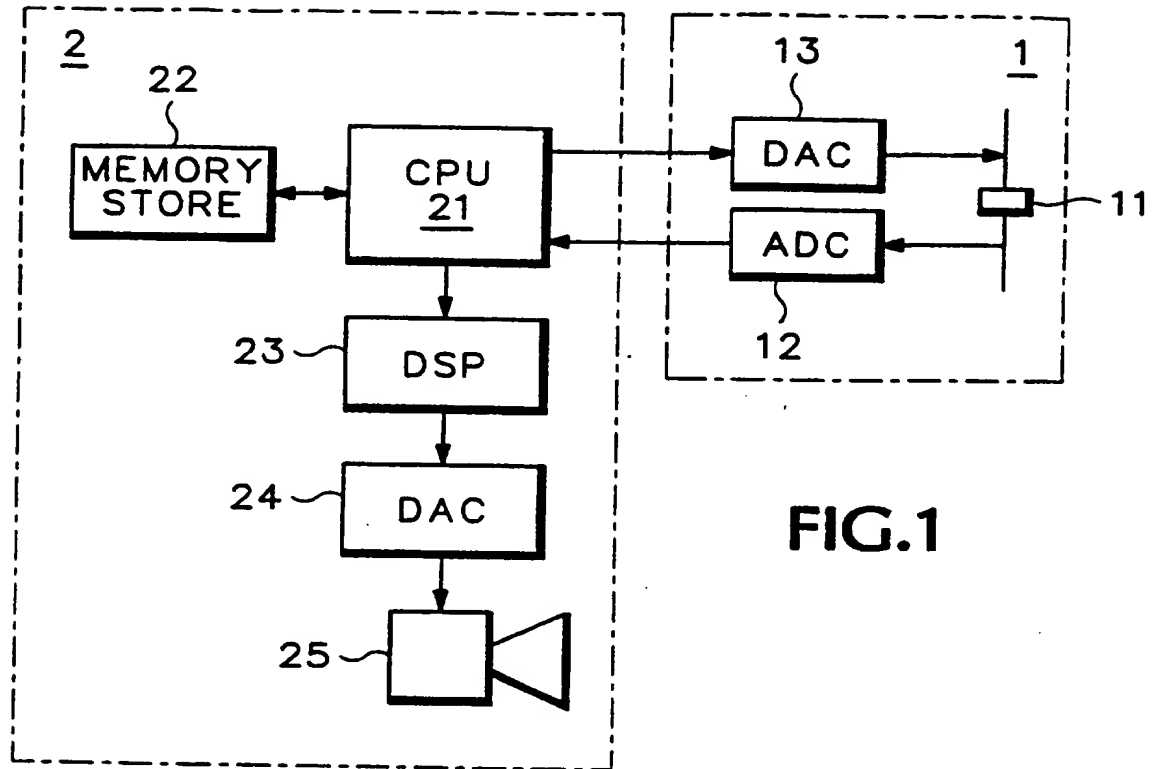
(57) An audio signal control system has a fader control 11, a memory store 22, a CPU 21 controlling operation of the memory store and outputting signals to a digital signal processor 23 and thence, via a DAC 24 to a speaker 25. The invention produces plural sequences of control parameters indicative of the position of the fader 11 and an identifier code representative of each sequence is stored in memory 22 to produce a first version. When a user modifies the position of the fader 11 so a sequence or sequences are modified and a second version of identifier code for the sequences is produced comprising the first version plus the modified sequence. With such a procedure, it is possible for a user to undo/redo versions of the fader control positions and thereby manipulate an audio signal.

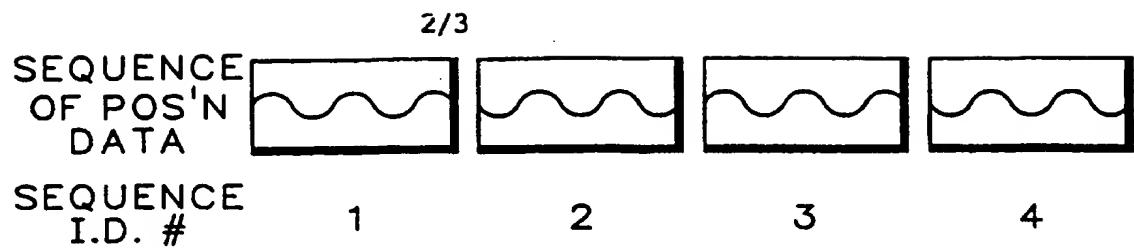


**FIG.1**

At least one drawing originally filed was informal and the print reproduced here is taken from a later filed formal copy.  
The claims were filed later than the filing date within the period prescribed by Rule 25(1) of the Patents Rules 1995  
This print takes account of replacement documents submitted after the date of filing to enable the application to comply with the formal requirements of the Patents Rules 1995

**GB 2 312 135 A**

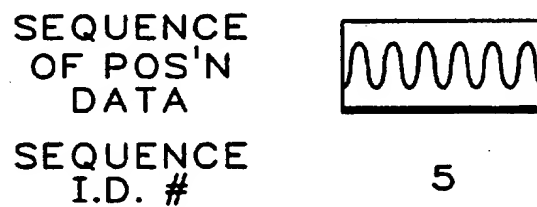




**FIG.3(a)**

VERSION 1 ( 1, 2, 3, 4, )

**FIG.3(b)**



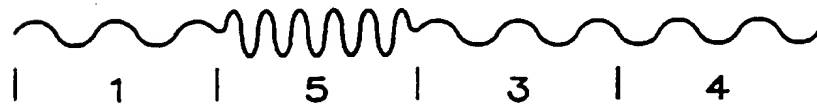
**FIG.3(c)**

VERSION 2 ( 1, 5, 3, 4, )

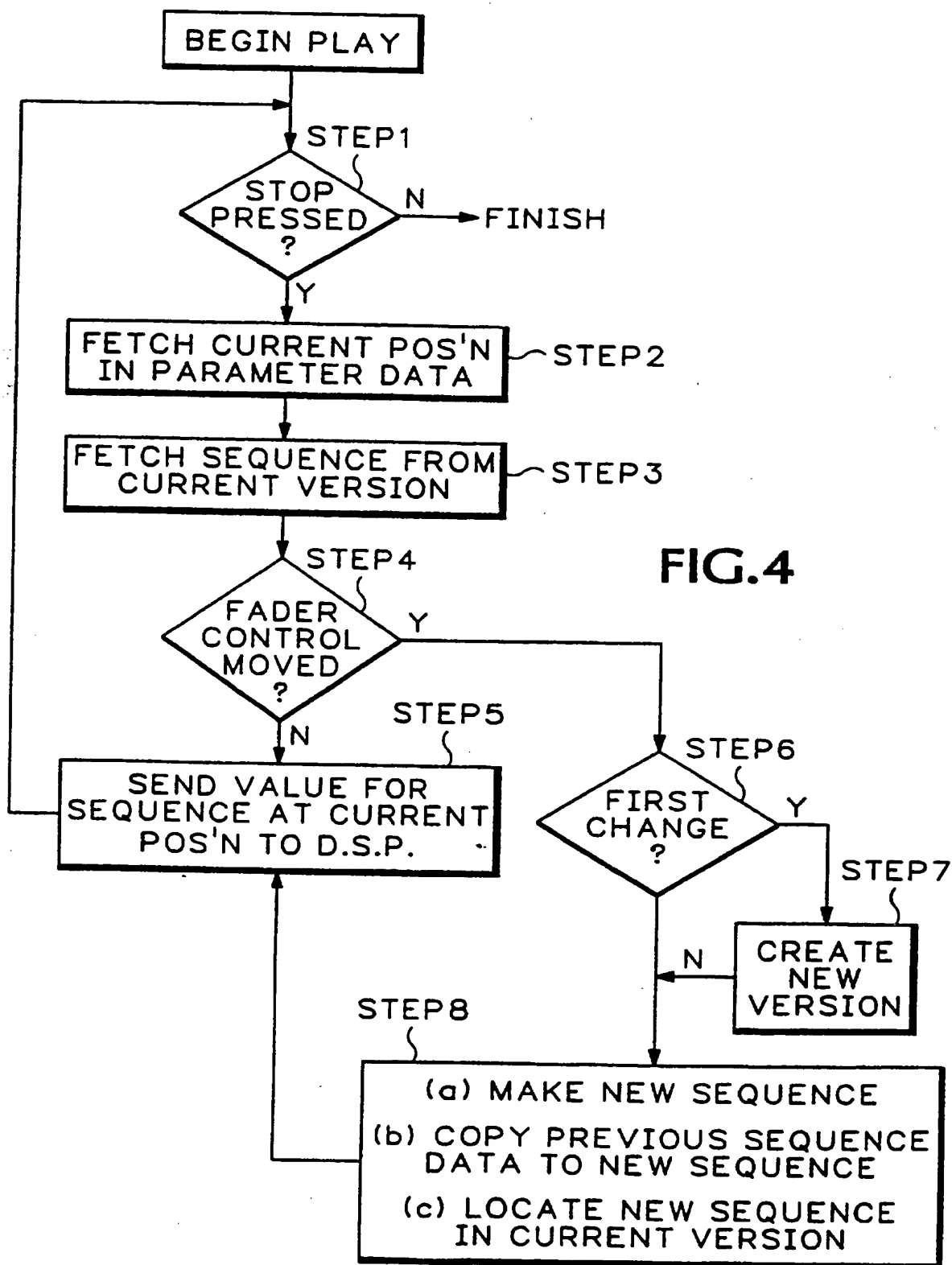
**FIG.3(d)**



**FIG.3(e)**



**FIG.3(f)**



AUDIO SIGNAL CONTROL SYSTEM AND METHOD THEREFOR

This invention relates to an audio signal control system and to a method of operating an audio signal control  
5 system.

When an audio engineer mixes several audio tracks together to form a final production version, the level of each channel for each track of the mix is varied throughout  
10 the production on a digital mixing console. Thus an audio signal is recorded via plural channels onto several audio tracks and each one of the audio tracks may be subsequently modified by an audio engineer. The modified audio signal is stored so that the final production audio signal is  
15 usually a result of several prior versions of the audio signal having been modified. In the digital mixing console each of the audio signal levels is defined by plural sequences of numeric values generated using a fader control coupled to an analog to digital converter (ADC).

20

The numeric values are stored and the levels are subsequently used by a digital signal processor (DSP) in conjunction with the associated digital audio samples in a mixing calculation, the result of which is converted to an  
25 analog signal in a digital to analog converter (DAC) and subsequently aurally played on a speaker.

Known audio production equipment may record the sequences of numeric values generated from the fader  
30 controls and store the sequences in memory, such as a disk drive.

When replaying the audio signal, the fader controls are moved approximately in synchronism in dependence upon  
35 the sequences stored in memory so that the fader controls mimic their original positions. When an audio engineer wishes to modify a track of the audio signal, the

appropriate fader control is moved to a new position and when the audio engineer is satisfied that the new position is correct then the new sequence of numeric values for the fader control is stored.

5

In the known equipment, in the step of storing the new version of the fader control numeric position, the previous version is overwritten.

10       The present invention seeks to provide an apparatus and a method in which current and previous versions of the numeric values generated by the fader control are stored so that an audio engineer may revert back to a previous version, if desired.

15

According to a first aspect of this invention there is provided an audio signal control system including means for determining plural sequences of a control parameter for generating an audio signal to produce a first version,  
20       means for storing said plural sequences of said control parameter, means for modifying said audio signal, and means for determining and storing only the sequence or sequences of said control parameter that were modified in modifying said audio signal, and means for reassembling said control  
25       parameter from said sequences derived from various versions of said sequence or sequences.

In a preferred embodiment, the means for modifying said audio signal produces a modified sequence or sequences  
30       of said control parameter to produce a further version of said sequence or sequences and means are provided for assembling the originally stored version of sequences altered by said modified sequence or sequences, whereby said original and said further version are stored by said  
35       storing means.

Preferably said means for storing is arranged to store N versions of said sequences and conveniently N may be equal to 10.

5 In a feature of this invention there is provided an audio signal control system including means for generating an audio signal connected to means for storing a control parameter associated with said audio signal, means for  
10 modifying said stored audio signal and at least a portion of said control parameter, said control parameter being divided into plural sequences and said portion being a sequence or sequences of said control parameter, means for storing at least one sequence of said control parameter that is modified by said modifying means whereby said means  
15 for storing stores said original control parameter and said modified control parameter.

Preferably the control parameter is produced by a position control means such as a slider control or a rotary  
20 control.

Advantageously means are provided for replaying said stored audio signal in dependence upon a desired serial sequence of said control parameter stored in different  
25 versions thereof, and said means may conveniently comprise a digital signal processor receiving signals from said storage means and outputting signals via a digital to analog converter to an audio speaker.

30 According to a further aspect of this invention there is provided an audio signal control method including the steps of determining plural sequences of a control parameter for generating an audio signal to produce a first version, storing said plural sequences of said control  
35 parameter as said first version, and when modifying said audio signal, determining and storing only the sequence or sequences that were modified to produce a second version

whereby both said first version and said second version are stored for subsequent recall, and assembling said control parameter from said sequences derived from said first version and said second version for utilisation.

5

In a currently preferred embodiment, the modified sequence or sequences produce a second version comprising said first version of sequences altered by said modified sequence or sequences whereby both said first version and  
10 said second version formed by said first version modified by said altered sequence or sequences are stored in their entirety. In another embodiment the modified sequence or sequences is stored by storing the change in parameter of the modified sequence or sequences rather than the actual  
15 modified sequence or sequences. In a further embodiment, only the modified sequence is stored with an appropriate address.

Preferably N versions may be stored and conveniently  
20 N may be equal to 10.

Advantageously each said sequence has a predetermined length, of for example 4096 bytes.

25 Preferably each sequence comprises parametric control data and a unique identifier code, which identifier code may be for example a numerical code. A version typically may comprise plural sequences plus associated identifier codes.

30

Conveniently, a predetermined sampling rate of for example 50Hz is used to determine the location of the sequence or sequences that are modified in a series string of such sequences.

35

It will be understood from the above that the system and method of this invention in the preferred embodiment



stores control parameters associated with an audio signal in a sequence of numbers so that a version of said control parameters may be stored. When the control parameter is changed to change the audio signal, the particular number  
5 of the sequence or sequences that is modified replaces the originally stored sequence (version) so that both original and modified versions of sequences are stored for subsequent use.

10 The invention will now be described by way of example with reference to the accompanying drawings in which:

Figure 1 shows a block schematic diagram of an audio signal control system in accordance with this invention,  
15

Figure 2 shows an entity relationship diagram in accordance with this invention,

Figure 3 shows in schematic form the manner by which  
20 numeric control data is split into blocks of data, each defining a sequence to produce plural versions of the control data, and

Figure 4 shows a flow diagram of the system and method  
25 in accordance with this invention.

The audio signal control system shown in Figure 1 has a mixing console 1 connected to a computer system 2. The mixing console is a digital device for manipulating and  
30 recording an audio signal. In the embodiment shown, the audio signal is modified by a slider, fader control 11 although it is to be understood that the mixing console may with modifications known per se may alternatively utilise rotary controls or pushbuttons (not shown). Further, the  
35 mixing console shown in Figure 1 depicts only a single fader control whereas in practice there will be plural fader controls 11. The fader control 11 generates a

control parameter indicative of the position of the fader control 11 along its movement track. The positional control parameter is an analogue signal which is converted to a digital signal by an analogue to digital converter 12 (ADC) which provides an output signal to a CPU 21. The CPU 21 has a memory 22 which may be in the form of a disk store. The memory 22 is used to store both the control parameter indicative of the position of the fader control 11 and the audio signal that is produced as a result of manipulation of the fader control 11. So as to play back the audio signal from the memory 22, the CPU 21 is connected to a digital signal processor (DSP) 23 which provides output via a digital to analog converter (DAC) 24 to an audio speaker 25. When replaying the audio signal the CPU provides output signals to a DAC 13 which controls the position of the fader control 11. In this respect the fader control 11 is usually motorised so that when the audio signal is being replayed the position of the fader control 11 mimics its original position when the audio signal was recorded. An audio engineer may modify the position of the fader control 11 so as to modify the positional control parameters that were originally recorded.

The control parameter of the fader control 11 that is stored is directly indicative of the audio signal that is recorded since the control parameter is derived from the position of the fader control along its track. The first version that is recorded of the control parameter (and hence the audio signal) is herein termed version 1. A modified version of the control parameter is herein termed version 2 and it is to be understood that plural modified versions may be recorded although for convenience it is expected that only about 10 versions will be stored, but such a number is not intended to be limitative.

Thus the control parameters that are stored may be representative for example of control of audio gain,

filtering, panning (i.e. left, right channel signal control for stereo and/or front, back channel control for quad) and muting. In this invention the control parameter is divided into plural sequences of blocks of data so that subsequent  
5 editing of a sequence or sequences results in an isolated change that is stored.

The entity relationship diagram shown in Figure 2 shows a sequence or block of recorded parametric control  
10 data 31. The block of data has a fixed storage size of, for example, 4096 bytes. Each sequence has a sequence identifier code 32 which may be simply a number, for example, 1, 107, or 563419. The sequence of the data together with the sequence identifier code is combined to  
15 form a sequence record 33 which is stored in a sequence store 34 which is a set of sequence records.

A version 35 is produced by a list of version identifier codes, e.g. 1, 107 or 563419. A control stream  
20 36 is a set of versions. Thus a control stream is a set of versions, one of which is the current, i.e. latest, version and each version is a list of sequence identifier codes. The identifier codes determine a sequence, i.e. block of control data.

25

Therefore, playing an audio signal is performed by considering the version to be played which uses the steps of looking up the record corresponding to identifier codes listed in the version and using the block of data in the  
30 sequence to adjust the position of the control fader 11. The sequences, i.e. blocks of data, are played back one by one.

At any point on a subsequent replay of the audio  
35 signal, the audio engineer may override the stored value from a sequence with a new value derived by the engineer moving the fader control 11 to a new position. When the

audio engineer is satisfied that it is at a desired position, the engineer causes the new position to be stored.

5        Thus, during replay of a current version and the control fader 11 position is altered, a new version of the control parameter is created. This version is a copy of the previous version but with a new sequence identifier code substituted in place of the sequence identifier code  
10 whose sequence, i.e. data block, was changed. The sequence identifier code therefore identifies the new block whose data contains the changed value. The original block remains unchanged as the version 1 and the modified version is then version 2. Subsequent changes throughout the same  
15 session are all part of the same (modified) version.

The above operation will now be described in greater detail with reference to Figures 3 and 4.

20        Referring to Figure 3, the control parameter, i.e. the location of the fader slider 11 is broken up into sequences of data each of block size 4096 bytes. In Figures 3(a) and 3(b), four sequences are shown each of 4096 bytes with the sequence parameter control data being shown in Figure 3(a)  
25 and the associated sequence identifier code being numerics 1, 2, 3, 4 respectively shown in Figure 3(b). The stored data is of the sequence identifier code so that the original control parameters for the fader control 1 (version 1) is 1, 2, 3, 4 (as shown in Figure 3(b)).

30

If now the audio engineer upon replaying the audio track decides to move the slider control 11 in the sequence identified by identifier code 2, i.e. the second block of data, to produce a sequence of positional data for that  
35 modified (second) position shown in Figure 3(c) then on that modified version 2 the block sequence identifier code is 1, 5, 3, 4 as shown in Figure 3(d). Thus the sequence

of positional data has changed from version 1, shown in the combined form of Figure 3(e), to the version 2 shown in Figure 3(f).

5        It will therefore be understood that the sequence identified by code 2 is used only in version 1 and the sequence identified by the code 5 is used only in version 2, with sequences identified by the codes 1, 3 and 5 being shared between versions 1 and 2. However it is only  
10 necessary to store the identifier codes 1, 3 and 4 once since they are shared between the two versions and the identifier code 2 for version 1 and the identifier code 5 for version 5.

15        The flow chart will now be described with reference to Figure 4 where at each time "play" begins, it is determined at step 1 if the control 11 is pressed to stop the audio signal. If "yes", the current position of the control data is measured in value samples in step 2. Thus if for  
20 example a sampling rate of 50Hz is used then after one minute, forty seconds, of sampling, the sample would be number 5000. Thus if the slider control is pressed by an audio engineer, the sampling is stopped and the sequence identifier code for the current position is determined.  
25 Thus with a block size of 4096 bytes and at a sample position of 5000 the sequence identity code is equal to the sample number divided by the block size. From such a calculation it is determined that the sequence identity code is equal to 2 remainder 904 (the block offset) at step  
30 3. The sequence (control data parameter representative of the fader slider position) associated with the sequence identifier code is read. If the audio engineer has not changed the sequence (indicated by a different position of the control 11) the original sequence (step 4) control  
35 parameter value is written to the DSP 23 at step 5 and the cycle repeated.

If however the fader slider control position has moved at step 4 then it is determined at step 6 if it is the first positional change during replay of version 1, i.e. the same playthrough. If at step 6 it is decided that it is the first change then a new version is created at step 7. At step 8, a new sequence, i.e. block of data, is written, the old block of data of version 1 up to the new sequence is copied and the new sequence is put into the current version. Thus in the example of Figure 3, version 2 is created by copying original version sequence 1, but with new sequence 5 substituting version 1, sequence 2.

The flow then steps to step 5.

If at step 6 it is determined that it is not the first change during the same playthrough and a new version has already been created then the flow moves to step 8. Therefore version 2 is created by copying original version 1 sequences 1, 3, 4 and substituting new version 2 sequence 5 for version 1 sequence 2.

At both steps 5 and 8, the data block offset is determined from the sample position (mod data block size), i.e. equals remainder of  $5000 \div 4096$  which block offset is written to the DSP.

Thus with the present invention a new version is made by copying the current version and writing a new block of data for a particular block that is altered into the version that is being played. A table of different versions will therefore be built up, any of which may be replayed and further modified.

The present invention therefore has the ability to "undo" a previous modification and is able to "redo" a previously modified version by appropriate selection of the version to be played.

It is currently envisaged that because it is only a numeric code identifier that is being stored for each sequence so the code identifiers for each sequence in a version will be stored. However it is envisaged that  
5 instead of storing the actual modified sequence it is the change in sequence (i.e. the  $\Delta$  sequence) that is stored. In another embodiment it is envisaged that instead of storing the code identifier for each sequence of a version, it is only the modified sequence that is stored for a  
10 subsequent version but in such a situation the user will have to identify which of the previous versions is to be associated with such a modified sequence.

by said altered sequence or sequences are stored in their entirety.

12. A method as claimed in claim 10, wherein the modified  
5 sequence or sequences is stored by storing the change in  
parameter of the modified sequence or sequences rather than  
the actual modified sequence or sequences.

13. A method as claimed in claim 10, wherein only the  
10 modified sequence is stored with an appropriate address.

14. A method as claimed in claim any of claims 10 -13,  
wherein N versions may be stored.

15 15. A method as claimed in claim 14, wherein N is equal to  
10.

16. A method as claimed in claims 10 - 15, wherein each  
said sequence has a predetermined length.  
20

17. A method as claimed in claims 10 - 16, wherein each  
sequence comprises parametric control data and a unique  
identifier code.

25 18. A method as claimed in claim 17, wherein said  
identifier is a numerical code.

19. A method as claimed in any of claims 10 - 18, wherein  
a predetermined sampling rate is used to determine the  
30 location of the sequence or sequences that are modified in  
a series string of such sequences.

20. An audio signal control system substantially as herein  
described with reference to and as shown in the  
accompanying Figures.

35 21. An audio rigid control method substantially as herein  
described with reference to and as shown in the  
accompanying Figures.





Application No: GB 9607571.8  
Claims searched: ALL

Examiner: Mr. SAT SATKURUNATH  
Date of search: 3 July 1997

**Patents Act 1977**  
**Search Report under Section 17**

**Databases searched:**

UK Patent Office collections, including GB, EP, WO & US patent specifications, in:

UK Cl (Ed.O): H4R: RSX, RSCB, RPNR, RPX; G5R: RB81

Int Cl (Ed.6): G11B, H04H

Other: Online: WPI, JAPIO, INSPEC

**Documents considered to be relevant:**

Category	Identity of document and relevant passage	Relevant to claims
X	GB 2306750 A QUANTEL - see especially figure 8 and pages 10,11	1, 5 and 10
A	GB 2258374 A SOLID - see especially figure 1 and pages 9-11	1, 5 and 10
A	US 4479240 McKINLEY - see especially figure 4	1, 5 and 10

X Document indicating lack of novelty or inventive step  
Y Document indicating lack of inventive step if combined with one or more other documents of same category.  
& Member of the same patent family

A Document indicating technological background and/or state of the art.  
P Document published on or after the declared priority date but before the filing date of this invention.  
E Patent document published on or after, but with priority date earlier than, the filing date of this application.

This Page Blank (uspto)